

**What Is Claimed Is:**

- 1           1.       An apparatus that provides a unified telephony solution,  
2 comprising:  
3           an application server configured to provide telephony services;  
4           a voice extensible markup language (VXML) browser configured to access  
5 telephony services through the application server;  
6           a telephony controller configured to access telephony services through the  
7 VXML browser; and  
8           a telephony gateway that provides an interface to a public switched  
9 telephone network (PSTN).
- 1           2.       The apparatus of claim 1, wherein the telephony controller includes  
2 a SIP framework with a SIP servlet container, wherein the SIP servlet container  
3 includes a plurality of SIP servlets for interfacing with a SIP network.
- 1           3.       The apparatus of claim 2, wherein a new telephony service can be  
2 added by including a new SIP servlet in the SIP servlet container.
- 1           4.       The apparatus of claim 2, wherein the plurality of SIP servlets are  
2 registered with a remote method invocation (RMI) registry.
- 1           5.       The apparatus of claim 2, wherein the telephony services provide at  
2 least two of:  
3           a telephone system;  
4           a call center;

5           an interactive voice response (IVR) system; and  
6           a voicemail system.

1           6.       The apparatus of claim 2, wherein the apparatus operates using a  
2   Voice Over Internet Protocol (VOIP).

1           7.       The apparatus of claim 2, wherein the application server is coupled  
2   to a database that provides access to the plurality of SIP servlets.

1           8.       A method that provides a unified telephony solution, comprising:  
2       receiving a request for a telephony service at a telephony controller; and  
3       in response to the request, accessing a telephony service provided by an  
4   application server;  
5       wherein the application server is accessed through a voice extensible  
6   markup language (VXML) browser;  
7       wherein performing the telephony service involves interfacing to a public  
8   switched telephone network (PSTN) through a telephony gateway.

1           9.       The method of claim 8, wherein the telephony controller includes a  
2   SIP framework with a SIP servlet container, wherein the SIP servlet container  
3   includes a plurality of SIP servlets for interfacing with a SIP network.

1           10.      The method of claim 9, wherein a new telephony service can be  
2   added by including a new SIP servlet in the SIP servlet container.

1           11.    The method of claim 9, wherein the plurality of SIP servlets are  
2 registered with a remote method invocation (RMI) registry.

1           12.    The method of claim 9, wherein the telephony services provide at  
2 least two of:

3           a telephone system;  
4           a call center;  
5           an interactive voice response (IVR) system; and  
6           a voicemail system.

1           13.    The method of claim 9, wherein the telephony services operate  
2 using the Voice Over Internet Protocol (VOIP).

1           14.    The method of claim 9, wherein the application server is coupled  
2 to a database that provides access to the plurality of SIP servlets.

1           15.    A computer-readable storage medium storing instructions that  
2 when executed by a computer cause the computer to perform a method that  
3 provides a unified telephony solution, the method comprising:  
4           receiving a request for a telephony service at a telephony controller; and  
5           in response to the request, accessing a telephony service provided by an  
6 application server through a voice extensible markup language (VXML) browser;  
7           wherein the telephony service involves interfacing to a public switched  
8 telephone network (PSTN) through a telephony gateway.

1           16.    The computer-readable storage medium of claim 15, wherein the  
2    telephony controller includes a SIP framework with a SIP servlet container,  
3    wherein the SIP servlet container includes a plurality of SIP servlets for  
4    interfacing with a SIP network.

1           17.    The computer-readable storage medium of claim 16, wherein a  
2    new telephony service can be added by including a new SIP servlet in the SIP  
3    servlet container..

1           18.    The computer-readable storage medium of claim 16, wherein the  
2    plurality of SIP servlets are registered with a remote method invocation (RMI)  
3    registry.

1           19.    The computer-readable storage medium of claim 16, wherein the  
2    telephony services provide at least two of:  
3           a telephone system;  
4           a call center;  
5           an interactive voice response (IVR) system; and  
6           a voicemail system.

1           20.    The computer-readable storage medium of claim 16, wherein the  
2    telephony services operate using the Voice Over Internet Protocol (VOIP).

1           21.    The computer-readable storage medium of claim 16, wherein the  
2    application server is coupled to a database that provides access to the plurality of  
3    SIP servlets.